

Advanced Troubleshooting Guide for ShoreWare CSTA Server

Version 1.0

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October 2008..



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Advanced Troubleshooting Guide for ShoreWare CSTA Server Integration

1. Introduction

This document assumes familiarity with the documentation describing product installation and configuration:

- *ShoreWare CSTA Server Installation*
- *ShoreWare OCS 2007/OCS 2007R2 Telephony Integration / ShoreWare LCS 2005 Telephony Integration*

Read these documents carefully as they provide enough information to find and fix the most common issues. This document is intended to be used as an advanced troubleshooting guide for administrators and support professionals with experience in troubleshooting Microsoft server environments. It suggests strategies and techniques for resolving problems with deployments of ShoreTel CSTA Server with a specific focus on Microsoft MOC/OCS 2007/OCS 2007R2.

Since there is no significant difference between MOC/OCS 2007/OCS 2007R2 and the MOC/LCS 2005, the 2005 versions are not explicitly mentioned here.

The following terms and abbreviations are used in this document.

Term	Definition
OCS	Microsoft Office Communications Server
MOC	Microsoft Office Communicator
CA	Connectivity Adaptor
LinkTSP	Link Tapi Service Provider
AD	Active Directory
DVM	Distributed Voicemail Server, also known as ShoreWare Remote Application Server
DVS	Distributed Voicemail Server, also known as ShoreWare Remote Application Server



2. The Environment

A lot of components are involved in the process of integrating the CSTA Telephony Capabilities.

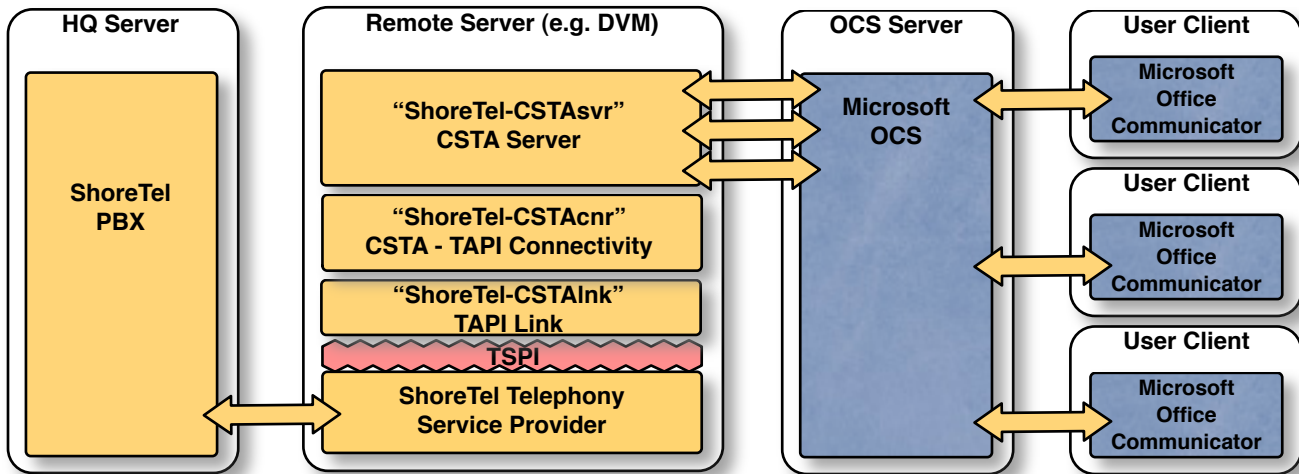


Figure 1 – Architecture

Each of these components needs to be properly configured and for each of these components it is possible that configuration settings lead to a failure

- ShoreTel System
- ShoreWare CSTA Server components (installed on a ShoreTel Remote Application Server also known as a Remote Server, DVM Server, or DVS)
- Microsoft Office Communications Server (running at the OCS Server)
- Microsoft Office Communicator Client (MOC, running on local workstations)

There are additional components - like Microsoft Active Directory or Microsoft SQL Server, Domain controllers etc., which are implicitly used. The proper configuration for these additional components is not covered by this document. This document focuses on the telephone Integration aspect only.

We are assuming that the MOC users are already configured for OCS usage in Active Directory.



3. Troubleshooting Strategy

The approach for narrowing down a phone integration issue depends on the context:

- Initial Phone Integration (“first time”) – does not work
- Existing Phone integration environment (partially) stopped working

3.1.Existing Phone Integration

Choose a top down approach:

- Check MOC settings regarding “Phone enabled”
- Check OCS settings regarding
 - Global settings like static routing, authorization
 - User specific settings like telURI, Phone Integration enabled
- Check CSTA settings regarding
 - General system availability
 - Licensing
 - Valid extension

3.2.Initial Phone Integration

Choose a bottom up approach

- Check CSTA settings regarding
 - General system availability
 - Licensing
 - Valid extension
- Check OCS settings regarding
 - Global settings like static routing, authorization
 - User specific settings like telURI, Phone Integration enabled
- Check MOC settings regarding “Phone enabled”

3.3.Ruling Out User Error—Dialing Formats

Users may complain that they are unable to dial certain numbers. Before commencing troubleshooting of the configuration, first confirm that the user is entering the numbers to be dialed correctly and that numbers entered into directories and databases being used are correctly formatted.

Numbers dialed by client applications such as Microsoft Office Communicator must be in a canonicalized “international” format, for example, +14085551234.



4. MOC Diagnostics

The MOC client knows three different states regarding the phone integration. These three states are immediately reflected by a visual feedback within the GUI

Normal State: phone integration is enabled and configured properly

In this case you will see a phone icon (1) and the info icon (2) does not show an error.

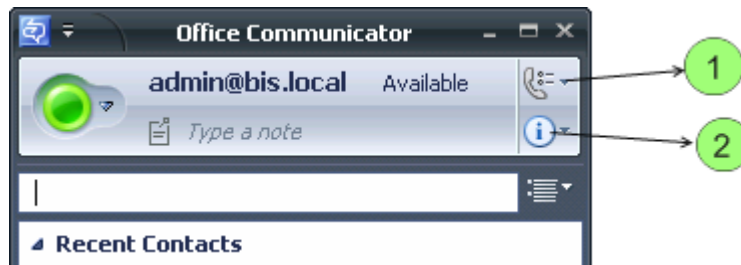


Figure 2—MOC is enabled and configured properly

Note: The MOC might show a phone icon indicating “Ok” (1) but also a Tray Icon with the contrary indication(2). After a while the “Phone System Error” within the Tray Icon will disappear automatically.

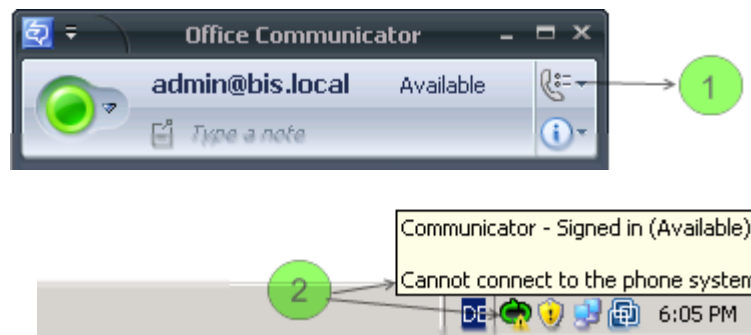


Figure 3—Spurious phone system error

Fail State I: Phone Integration Not Enabled

In this case the MOC does not show an error, but there is no phone icon

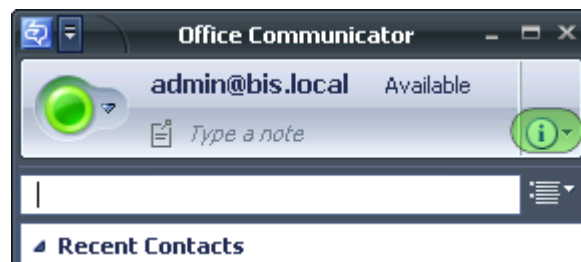


Figure 4—Phone integration not enabled

Fail State II: Phone Integration Enabled but Phone System Error Occurred

In this case the MOC shows an error via the **info icon** and the **phone icon** is missing



Figure 5—Phone integration is enabled but an error has occurred

More information about the error is displayed via the **info icon**.

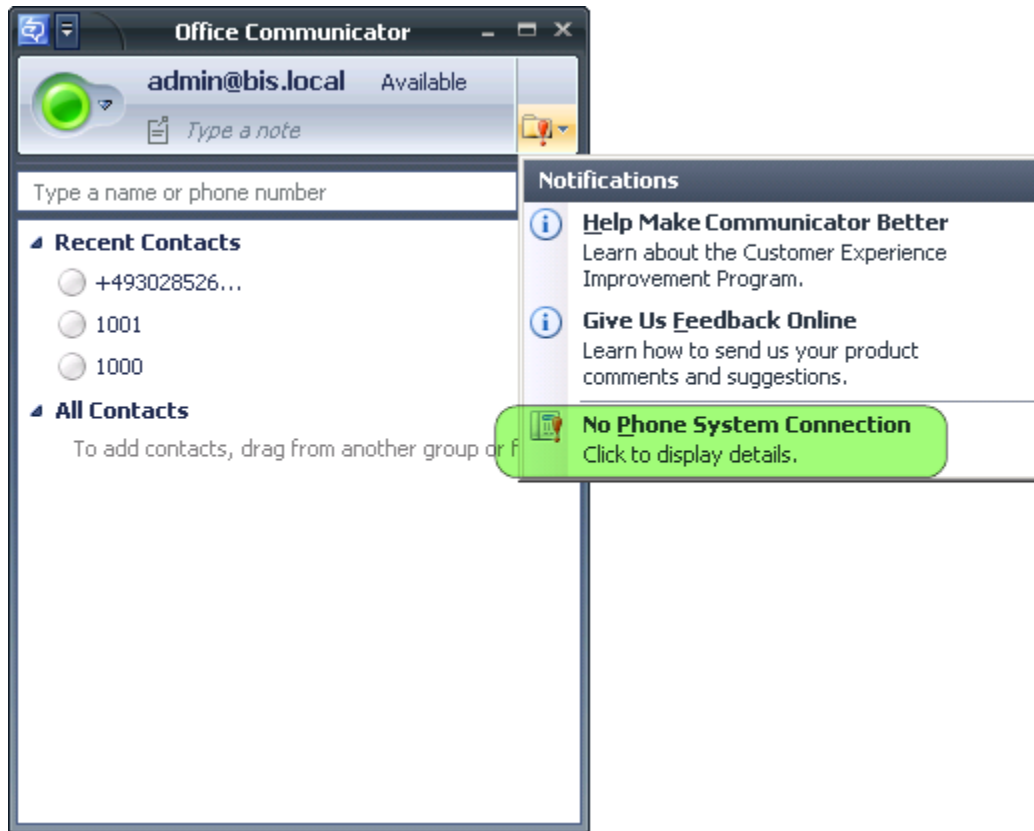


Figure 6—Phone integration error notification

These two fail states (“Fail State I” and “Fail State II”) are discussed below in separate sections.



4.1. Telephony Integration Not Enabled (Fail State I)

Symptoms of this scenario are:

- **Phone icon** is not present
- **Info icon** indicates no errors
- **Phone Integration enabling checkbox** is disabled (“grayed out”)

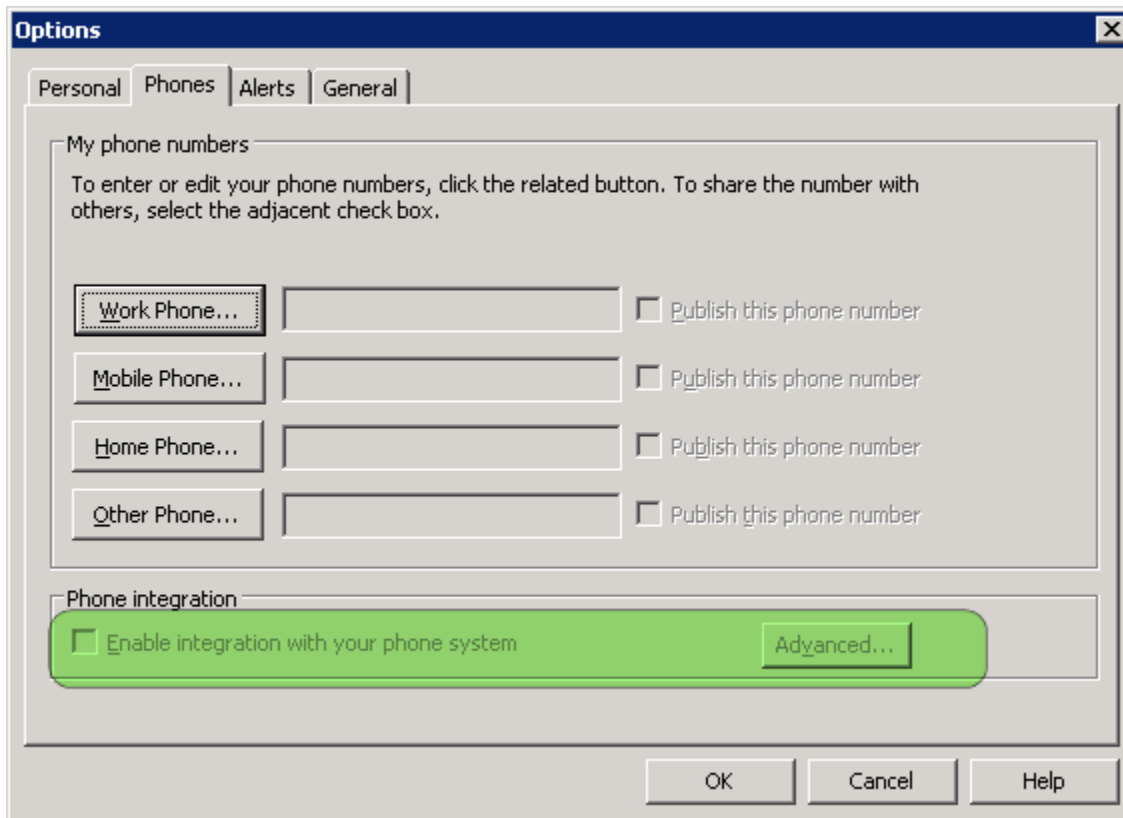


Figure 7—Phone Integration checkbox is disabled

Cause: PBX integration has not been enabled for this user on the OCS server

Solution: Check the OCS configuration for this user and enable the PBX integration

Cause: MOC 2005 running on a system that is missing certain registry keys

Solution: The MOC 2005 client needs certain registry keys to be set before it will enable the phone integration. The registry keys are:

```
[HKEY_LOCAL_MACHINE\SOFTWARE\Policies\Microsoft\Communicator]

"EnablePhoneControl"=dword:00000001

"EnableConferencingService"=dword:00000001

"EnablePC2Phone"=dword:00000001
```



4.2.No Phone System Connection (Fail State II)

Symptoms of this scenario are:

- **Phone icon** is not present
- **Info icon** indicates a failure (“No Phone System Connection”)

Cause: OCS Environment not properly configured

Solution: Check the configuration of the OCS server, the ShoreWare CSTA server, and the PBX.. If you are an Administrator then you could enable the MOC logging feature. Follow the instructions below for *CSTA diagnostics*. If everything seems to be OK then move on to *OCS diagnostics*.



5. CSTA Diagnostics

The ShoreWare CSTA Server consists of three subcomponents

- ShoreTel-CSTAsvr – the CSTA Server
- ShoreTel-CSTAcnr – Connectivity Adaptor
- ShoreTel-CSTAlnk – TAPI bridge

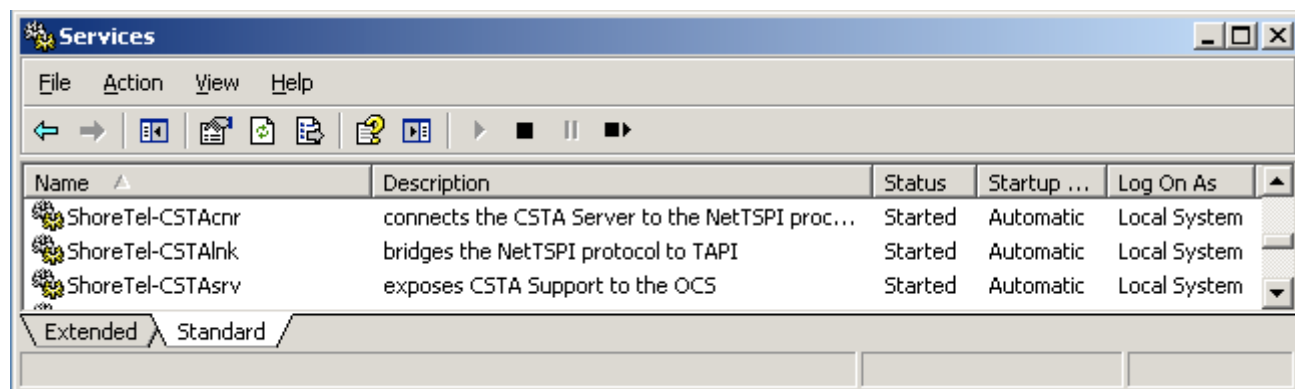


Figure 8—Windows Services Panel

Each of these software components runs as a distinct Microsoft Windows System Service so first check to be sure that all of these services are up and running (this is explained below).

Note: The term “CSTA Server” generally refers to the top level component which is the ShoreTel-CSTAsvr system service. The other two components (ShoreTel-CSTAcnr and ShoreTel-CSTAlnk) do not expose “public capabilities” and are only used internally by the ShoreTel-CSTAsvr component.

5.1. Diagnostic Session Facilities

The CSTA Server has two built-in diagnostics session facilities. Both of them are accessible using a standard Telnet program such as the one included with the Microsoft Windows operating systems.

Note: In the examples of Telnet use in this manual the address of the server is specified as the loopback address **localhost** to illustrate Telnet running on the same host as the CSTA server. Telnet could be running on an computer with network access to the CSTA server which would then be identified by its IP address.

Note: These examples also use the default value of the CSTA server’s port number which is 26535. A different port number can be configured for the CSTA server by changing the appropriate entry in the CSTA server’s *Default.conf* configuration file. (This is described later in this manual.)

The SuperVisor Session

The SuperVisor session can be used to check licensing, PBX reach-ability, and version information.

```
telnet localhost 26535
SuperVisor
BYE
```

(connects to the CSTA Server)
(login to a SuperVisor session)
(ends a SuperVisor sessions)



The STLI Session

The STLI session can be used to check basic telephony features (e.g. device monitors).

```
telnet localhost 26535
STLI
BYE
```

(connects to the CSTA Server)

(login to a STLI session)

(ends an STLI sessions)

Note: The Telnet client has some restrictions:

- special keys do not work, e.g. if you have a type a command you cannot use the backspace key for corrections. (That's because the backspace character will be part of the command).
- There is no initial prompt. You have to start your diagnostic session by blindly typing the appropriate command (**STLI** or **SuperVisor**)

5.2.CSTA Server Service(s) Not Running

The CSTA Server consists of three sub components which run as Windows System Services.

These three services must be up and running. Use the Microsoft Windows Services Panel or the ShoreWare Director Quick Look Distributed Server Maintenance Page to check the status of these services.

ShoreTel
ShoreWare Director
Build 14.5.5403.0
Logoff Juan Rubio

Administration

- Users...
- Trunks...
- IP Phones...
- Switches...
- Call Control...
- Voice Mail...
- Auto-Attendant
- Menus
- Workgroups
- Schedules
- Call Manager...
- System Directory
- Application Servers
- Conference Bridges
- IM Servers
- Sites
- System Parameters...
- Preferences

Maintenance

- Quick Look
- Voice Mail Servers
- Switch Connectivity
- Conference Ports
- Event Log...
- Services
- Event Filters

Reporting

Distributed Server Maintenance Last updated: 2/25/2009 3:19:00 PM (GMT -08:00) [refresh](#)
on 10.10.0.10 Local time: 2/25/2009 3:19:40 PM (GMT -08:00) [Help](#)

[Go to Quick Look Page](#)

Status

Name	Status
Communication	OK
TAPI	OK
Notification	OK
SMTP Send	OK

Services [Go to All Services](#)

Service	Description	Status	Command
ShoreTel-CSISSVC	ShoreWare CSIS Server	Running	Select...
ShoreTel-CSTAacnr	ShoreTel-CSTAacnr	Running	Select...
ShoreTel-CSTAInk	ShoreTel-CSTAInk	Running	Select...
ShoreTel-CSTAAsrv	ShoreTel-CSTAAsrv	Running	Select...
ShoreTel-DBUpdateSvc	ShoreWare Remote DBUpdate Service	Running	Select...
ShoreTel-DRS	ShoreWare Distributed Routing Service	Running	Select...
ShoreTel-DTASvc	ShoreWare Distributed Telephony Application Service	Running	Select...
ShoreTel-EventSvc	ShoreWare Event Service	Running	Select...
ShoreTel-EventWatch	ShoreWare Event Watch Server	Running	Select...
ShoreTel-IPCS	ShoreWare IP-Phone Configuration Service	Running	Select...
ShoreTel-IPDS	ShoreWare IP-Phone Display Service	Running	Select...
ShoreTel-MailServ	ShoreWare Voice Mail Message Server	Running	Select...
ShoreTel-PortMgr	ShoreWare Voice Mail Port Manager	Running	Select...
ShoreTel-RemoteLoginSvc	ShoreWare Remote Login Service	Running	Select...

Figure 9— ShoreWare Director Quick Look Distributed Server Maintenance Page



If these services must be manually restarted you should use the following sequence for start up:

1. ShoreTel-CSTAlnk
2. ShoreTel-CSTAcnr
3. ShoreTel-CSTAsvr

This sequence is recommended but not strictly required since all of these services have built-in automatic synchronizing and recovery capabilities but the use of this sequence will decrease the time which is needed to synchronize between these components and the availability of the whole system is reached much faster.

5.3. Invalid Device Extension (Monitor)

Login via STLI session.

```
telnet localhost 26535
STLI
```

(connects to the CSTA Server)
(login to a STLI session)

Assuming that you want to use the device 3000, than you could use the **MonitorStart** command,

```
MonitorStart 3000
```

(starts monitor on x3000)

If the MonitorStart command responds with an error indicating an `invalidMonitorObject` then you are using an extension number that does not exist or is invalid:

```
error_ind UniversalFailure operationalError
invalidMonitorObject
```

In the example above, an attempt is made to start a monitor on extension 3000 but the `invalidMonitorObject` reference in the response indicates that there is no extension 3000 that can be monitored on the PBX.

5.4. Wrong Listener Port Configuration

Using Telnet on the DVM host (that is the computer where the CSTA Server is running) enter the following (note that the characters **SuperVisor** that you type will not be echoed back on the screen—you'll be typing "blind"):

```
telnet localhost 26535
SuperVisor
error_ind SUCCESS SuperVisor
BYE
```

(connects to the CSTA Server)
(login to a SuperVisor session)
(response indicates success)
(disconnects from session)

If an error message such as `connection refused` or something similar appears, the port is probably not 26535. Please check the configuration. The login port is configured during the installation process and may be manually configured via the setting **loginPort** inside the *Default.conf* configuration file.



5.5.Listener Port Already In Use by Another Process

Using Telnet on the DVM host (that is the computer where the CSTA Server is running)

Type the following (again, the word **SuperVisor** that you type will not be echoed back on the screen).

```
telnet localhost 26535
SuperVisor
```

If the response is neither `connection refused` nor `error_ind SUCCESS SuperVisor` then it is very likely that another process is listening on 26535. To confirm this, shutdown the **ShoreTel-svr** process and try this Telnet test again. If the same response is received, then you are sure that another process is using port 26535.

Solution: Either shutdown the foreign process, or if this is not possible, choose a different listener port for the CSTA server via the **loginPort** setting inside the *Default.conf* file and restart the **ShoreTel-svr** process.

Note: We recommend that you use Netstat, a built-in Microsoft Windows Networking tool which is capable of showing the listeners and the the related program binaries:

```
C: netstat -a -b
```

This might help you to identify the foreign process which occupies the 26535 port. Be aware that the netstat command runs extremely slowly in conjunction with the `-b` parameter.

5.6.Requests from the OCS Host are Blocked

Login to the OCS host (that is the computer where the OCS2007 Server is running). Lets assume that the DVM host (that is where the CSTA Server is running) has the IP-Address 10.99.0.10.Type:

```
telnet 10.99.0.10 26535
SuperVisor
```

(Note that the word **SuperVisor** that you type will not be echoed back on the screen—you'll be typing "blind".) The SuperVisor login command should respond with the message:

```
error_ind SUCCESS SuperVisor
```

If you are seeing a connection refused error message or something similar, it is very likely that there is a network connectivity issue or a firewall preventing connections from the OCS host to the DVM host for the specified port.



Solution: Ping end to end, then verify with your network administrator that there aren't any network rules or policies that prevent access. Change firewall rules to allow TCP connections from the OCS host to the DVM host for the specified port.

5.7. License expired

Use the **License** command of a SuperVisor Telnet session to check the license.

This command responds with information about the installed license.

```
License
error_ind SUCCESS License
maxMonitors: 9
days left: 60
expired: no
```

If the license command indicates an expired license please contact your ShoreTel sales representative to obtain a new license file. Install a valid license for the CSTA Server by copying and pasting it into the CSTA Server *Default.conf* file or by using the **License Tool** application.

5.8. Monitor Points Exceeded

To check how many monitor points are actually in use, utilize the **ShowDeviceMonitors** command after starting a SuperVisor session using Telnet. This command lists all the monitored devices (each of them consuming a single license)

```
ShowDeviceMonitors
error_ind SUCCESS ShowDeviceMonitors
1000 sticky:false active:true observers:1
1001 sticky:false active:true observers:2
1002 sticky:false active:true observers:1
1003 sticky:false active:true observers:1
1004 sticky:false active:true observers:5
1005 sticky:false active:true observers:1
1006 sticky:false active:true observers:1
1007 sticky:false active:true observers:1
```

When a device is already monitored and another monitoring request for the same device is received by the CSTA Server it does not consume a new license. The number of active monitors for a given extension is indicated by the *observers* value. In this example there are 8 used monitors (1000-1007, requiring at least 8 licenses) however there are a total of 13 active monitors because there are 5 observers on extension 1004 and 2 observers on extension 1001.



If you suspect that the number of monitored devices allowed by your license has been exceeded you can directly attempt to start monitoring a device that is not yet being monitored using an STLI session. For example:

```
telnet localhost 26535
STLI
error_ind SUCCESS STLI
MonitorStart 1000
```

If the **MonitorStart** command responds with an error that references `LICENSEERROR MONITORINGPOINTSEXCEEDED`, it means that you do not have a license with permission for enough monitored devices.

```
error_ind LICENSEERROR MONITORINGPOINTSEXCEEDED
Monitor not set
```

Solution: Obtain a new license file with a larger monitor limit. Replace the existing license with the larger license by copying and pasting it into the CSTA Server *Default.conf* file or by using the **License Tool** application.

5.9. Server Port Cascading Failure

ShoreWare CSTA server functionality is based on three separate subcomponents. These components are cascaded together via TCP/IP sockets.

Usually you do not have to change any of these ports. The Installer chains up these ports correctly using the default values as shown in Figure 10.

Figure 10 shows the keyword entries in the configuration files for each component and the default values. Installations should reserve the default ports of 26000, 26001, and 26535 for the ShoreWare CSTA Server. Only if it is essential that another process use one of the default ports, should the default port settings in the configuration files be overridden.

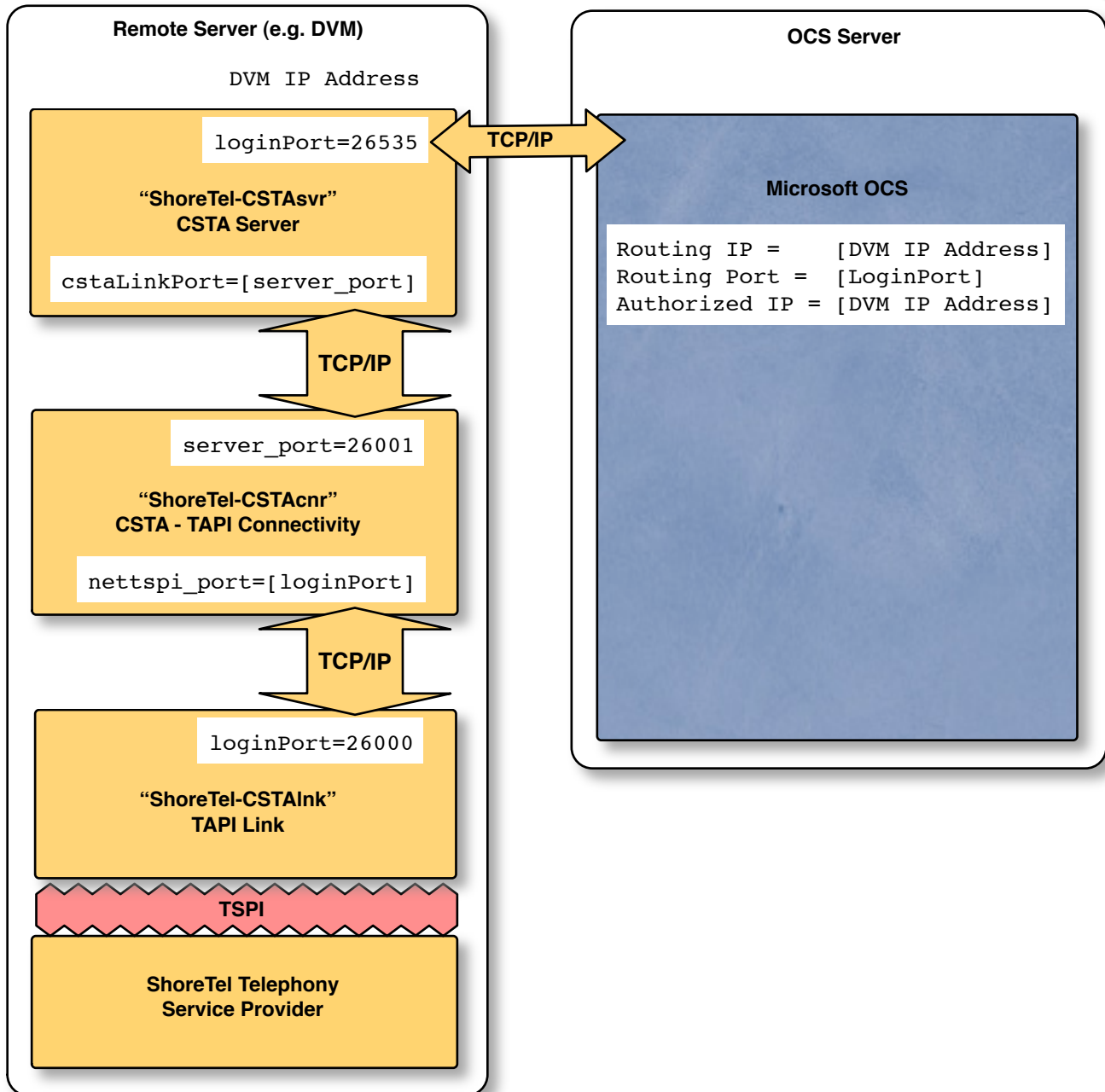


Figure 10—Cascading TCP ports



5.10. Configuration Files

By default, the configuration files for the CSTA Server components are stored in the directory *C:\Program Files\Shoreline Communications\ShoreWare Server\CSTA\Config*

Note: You do not usually need to modify these files as these are already properly configured by the CSTA Server Installer.

Note: As a best practice, you should always create a backup copy of a configuration file before modifying it.

ShoreTel-CSTAsvr Settings

The configuration file for the ShoreTel-CSTAsvr is named *Default.conf*. It contains the following keys.

Key	Default	Comment
loginPort	26535	Listener Port. This port is to be used in the OCS routing settings
license		Default is empty: demo mode with 1 monitor, Otherwise: a hexdump on a single line with encrypted licensing data
logFileMaxSize	100	Maximum size of a log file In KByte units. Actual size is limited only by file system constraints. The recommended range is 10 to 2000.
logFileMaxBackups	2	Number of log files to keep during log rotation. Actual number is limited only by file system constraints. The recommended range is 0 to 30.
debugLevel	0	0 = debug logging switched off 9 = for logging during debug sessions
cstaLogEnabled	0	Write csta.log (0 or 1)
interfaceLogEnabled	0	Write Interface.log (0 or 1)
cstaLinkAddress	127.0.0.1	The ip address for the host where ShoreTel-CSTAcnr service is running. By default this is the loopback address because it is on the same host as the ShoreTel-CSTAsvr service.
cstaLinkPort	26001	The listener port of the ShoreTel-CSTAcnr. Note: this port is explicitly set to 26001 during the installation
onNetPrefixList		List of prefixes used if ShoreTel On-Net Dialing feature is activated. All prefixes are listed on a single line separated by semicolons (";") By default this is empty. Note: Configuring the CSTA server for On-Net Dialing also requires making corresponding entries in the <i>Dialplan.conf</i> file (see below).
maxExtensionLength	5	Maximum number of digits in an extension number. Should be set to 3, 4, or 5.



rfc2806PrivateContext		"Private Context" string used to construct RFC2806 telURIs for extension numbers that have no corresponding DID number. The DID Prefix entered during installation is unique to the PBX and is used for this purpose.
setsignalhandler	1	For debugging only in the case a crash occurs: 0 : don't create a coredump 1 : create a coredump

ShoreTel-CSTAcnr Settings

The configuration file for the ShoreTel-CSTAcnr is named *ca_config.properties*. It contains the following keys.

Key	Default	Comment
server_port	26001	The listener port.
nettspi_hostname	127.0.0.1	The ip address for the host where the ShoreTel-CSTAlnk service is running. By default this is the loopback address because it is on the same host as the ShoreTel-CSTAcnr service.
nettspi_port	20000	The port of the ShoreTel-CSTAlnk service.

ShoreTel-CSTAlnk Settings

The configuration file for the ShoreTel-CSTAlnk is named *LINKTSP.INI*. It contains the following keys.

Key	Default	Comment
loginPort	20000	The listener port.
Driver	RpcTspX.tsp	The TAPI TSP driver DLL

ShoreTel-CSTAcnr and ShoreTel-CSTAlnk Logging Settings

The ShoreTel-CSTAcnr and ShoreTel-CSTAlnk share a common configuration file that controls logging behavior named *log4j-config.xml*. This is an XML file that contains additional key value pairs using XML tags. The key "level value" appears once for each component. To change the level of detail in the log files produced, change the value for these keys.

Key	Default	Comment
level value	ERROR	Logging level. The default, "ERROR", will log only errors. To capture all debugging information, change this value to "DEBUG". To stop all logging change the value to "OFF".



Dial Plan Settings for uaCSTA Clients

The ShoreWare CSTA Server provides a telephone number mapping function for clients using the uaCSTA protocol such as Microsoft OCS software. Mapping, or translation, of telephone numbers is controlled by the *DialPlan.conf* configuration file. (The *Default.conf* file provides extension length and On-Net Dialing prefix values.) *DialPlan.conf* contains two tables each with two columns of values. The first columns of both tables start in the first character position and the second columns start after a run of tabs and/or spaces.

The first table is known as the “Extension Translation Table” and it starts after a line that reads:

```
--Extension Translation Table
```

The entries in this table tell the gateway how to map references to internal extensions between DID numbers and extension numbers that can be dialed by the PBX. The first column contains characters to match with an external number (including country code). The second column contains digits to substitute in order to form the corresponding extension number.

The second table is known as the “External Translation Table” and it starts after a line that reads:

```
--External Translation Table
```

The entries in this table tell the gateway how to map references to external numbers between DID numbers and digit sequences that can be dialed by the PBX. The first column contains characters to match with an external number (including country code). The second column contains digits to substitute in order to form the corresponding dialable sequence.

Note: All entries are evaluated in the order they appear in the *DialPlan.conf* file. For translation to occur correctly, make sure that every entry is in the correct order: from most specific to least specific.

Translating Numbers Received from uaCSTA Clients

Telephone numbers received from a uaCSTA client are first examined to determine if they are already dialable extension numbers (based on their length) or On-Net Dialing extensions (based on their prefix). If so, no translation is required.

Otherwise, these numbers are processed by matching the leading characters of the telephone number provided with an entry in the first column of the Extension Translation Table. If a match is found, the matching portion of the provided telephone number is replaced with the digits found in the corresponding entry in the second column.

The examples below illustrate the substitution process for mapping to extension numbers:

First Column	Second Column	Number Provided	Resulting Extension
+14085551	11	+14085551234	11234
+14085551	589	+14085551234	589234
+140828512	7	+14085551234	734
+14085551234	456	+14085551234	456



If no match is found in the Extension Translation Table, then the matching process is repeated using the External Translation Table. The leading characters of the provided telephone number are matched with an entry in the first column of the External Translation Table. If a match is found the matching portion of the provided telephone number is replaced with the digits found in the corresponding entry in the second column.

These examples illustrate the substitution process for mapping to a dialing sequence for an external number:

First Column	Second Column	Number Provided	Resulting External Dialing Sequence
+	9+	+14085551234	9+14085551234
+1	91	+14085551234	914085551234
+1408	9	+14085551234	95551234
+1212	81212	+12125551234	812125551234

Translating Numbers to be Delivered to uaCSTA Clients

Telephone numbers received from the PBX are first examined to determine if they are extension numbers (based on their length) or On-Net Dialing extensions (based on their prefix). If so, numbers are converted from extension numbers to numbers in “international format” by matching the leading digits of the extension number with an entry in the second column of the Extension Translation Table. If a match is found the matching portion of the extension number is replaced with the digits found in the corresponding entry in the first column. If no match is found no translation is performed. If all the digits in the second column match the number in question, then all the digits are replaced with the entry found in the corresponding first column.

These examples illustrate the substitution process:

First Column	Second Column	Extension Number	Resulting Number
+14085551	11	11987	+14085551987
+14085551	589	589123	+14085551123
+140828512	7	789	+14082851289
+14085551234	456	456	+14085551234

Telephone numbers received from the PBX that correspond to external numbers are converted to DID numbers by matching the leading digits of the external number with an entry in the second column of the External Translation Table. If a match is found the matching portion of the number is replaced with the digits found in the corresponding entry in the first column. If no match is found then no translation is performed. If all the digits in the second column match the external number, then all the digits are replaced with the entry found in the corresponding first column.

These examples illustrate the substitution process:

First Column	Second Column	External Number	Resulting Number
+1	1	14085551987	+14085551987
+1		4085551987	+14085551987

Configuring Dialplan for On-Net Dialing

If the On-Net Dialing feature is enabled additional entries in the Extension Translation Table. As described above, the CSTA Server determines if a number is an extension number or an external



number based on its length. Telephone numbers too long to be extension numbers are considered external numbers unless they start with one of the On-Net Dialing prefixes configured in the *Default.conf* configuration file (see above).

If On-Net Dialing is configured then the extension numbers used in the mapping process will consist of the On-Net Dialing prefix followed by an extension number of the standard length and the Extension Translation Table must contain appropriate entries.

For example, a PBX is configured to use 3 digit extensions and an On-Net Dialing prefix of 589. All extensions associated with that prefix are associated with the DID range (408)555-1xxx. In this case the On-Net Dialing prefix 589 would be added to the list of On-Net Dialing prefixes configured in the *Default.conf* file and there would be an entry in the Extension Translation Table of the *Dialplan.conf* file with **+14085551** in the first column and **589** in the second column. This example is also illustrated in the second rows of each of the two sets of Extension Translation Table examples above.

5.11.Installer Parameters

During installation of the CSTA Server the installer requests two parameters: a **Trunk Access Code** and a **DID Prefix**. These parameters are used by the installer when it generates an initial *DialPlan.conf* file and create default entries in both the Extension Translation Table and the External Translation Table.

The DID Prefix consists of the unique digits, including the country code and the first digit of the extension range, that define a DID range in the PSTN. For example, if the DID range is +1(408)555-7000 through +1(408)555-7999 it means that the carrier has allocated all PSTN numbers starting with the digit sequence 14085557 to the PBX in question.

The DID prefix that is requested is also mapped internally to the following configurations key and stored into the *Default.conf* file.

Parameter Label	Internal Parameter Name	Comment
DID Prefix	rfc2806PrivateContext	MOC 2007 requires that extensions that cannot be mapped to any number in the PSTN must be specially encoded when placed in the uaCSTA protocol. This encoding is specified in RFC2806 which requires a "Private Context" string which uniquely identifies the PBX. The DID prefix for the PBX satisfies this requirement and is used for this purpose by default.

In a typical configuration the PBX with the DID prefix 14085557 would have these DID numbers assigned to extensions 7000 through 7999 respectively. The installer supports this basic configuration by creating a single entry in the *DialPlan.conf* file that maps PSTN number +1(408)555-7xxx to extension 7xxx. The entry in the Extension Translation Table in *DialPlan.conf* for this example is:

```
+14085557      7
```

This entry tells the CSTA Server to map to internal extension numbers by replacing the leading digits "+14085557" with just the digit "7" and to convert from internal extension numbers by replacing the leading digit "7" with the digits "+14085557".



Note: The installer only asks for a single DID Prefix so the one entered should be the primary DID prefix associated with the PBX. Additional DID range mappings must be added to the *DialPlan.conf* file manually. The installer assumes that the PBX conforms to the convention that all extensions corresponding to the DID range start with the last digit of the ID Prefix. If this is not the case, the *DialPlan.conf* file **must** be manually corrected after the installation process is completed. ShoreTel Professional Services may be engaged to assist with this process.

The installer uses the trunk access code parameter to create a default entry in the External Translation Table for converting canonical numbers to a digit sequence that can be dialed. It does this by creating an entry that prefixes canonical numbers with the trunk access code.

For example, if the trunk access code entered was “9” then the entry created in the External Translation Table in *DialPlan.conf* would be as follows:

```
+          9+
```

Note: The installer does not create default entries in the External Translation Table for mapping external numbers. In the basic configuration external numbers are delivered to the uaCSTA client as they are received from the PBX. If mapping of these numbers is required, appropriate entries must be created in the *DialPlan.conf* External Translation Table.

The example below illustrates the importance of entering the correct DID Prefix. In this case a user has extension 1100 and their OCS telURI setting is +14085551100.

DID Prefix	Resulting extension	comment
+14085551	1100	ok
+1408555	51100	wrong
+140828511	100	wrong
+4930828526	+14085551100	wrong

Troubleshooting DialPlan.conf

To confirm an issue with the *DialPlan.conf* file

1. Prepare a list of valid extensions and their corresponding telURIs as entered into OCS settings
2. Open the *DialPlan.conf* file with a text editor
3. For each telURI, scan the first column in *DialPlan.conf* and find the **first** entry that matches the leading digits of the telURI then replace the matching digits with the digits in the second column and match the result against the correct extension.
4. For each telURI, scan the second column in *DialPlan.conf* and find the **first** entry that matches the leading digits of the extension then replace the matching digits with the digits in the first column and match the result against the correct telURI.

If any converted number doesn't match, confirm that the telURI entered into OCS was valid. If so, the *DialPlan.conf* file is the source of the problem.

Solution: Manually correct the *DialPlan.conf* file by reordering or correcting existing entries, deleting invalid entries and/or adding missing entries then restart the CSTA server components.



6. OCS Diagnostics

The majority of phone integration failures result from mistakes in OCS configuration. Always begin with a simple check of the following:

- Host Routing and authorization settings (which are global settings)
- Ip address and port correct ?
- Replace URI checkbox enabled ?
- Matching routing table rule correct ?
- User specific phone settings (which are – as the name implies – personal settings)
- Is the telURI entered correctly ?

If this check does not lead to a solution than the usage of the OCS logging tool is to be considered.

Note: OCS logging has to be explicitly enabled. The examples in this illustration all refer to an OCS log file generated as documented in *Section 8.1 OCS Logging*, below.

6.1.No Matching Routing Table Rule

Symptom: The message flow in the logfile is similar to:

```
Text: Non-trusted source sent an FQDN/IP that doesn't match a
routing table rule

Result-Code: 0xc3e93c5e SIPPROXY_E_ROUTING

SIP-Start-Line: INVITE sip:callcontrol@dvm1.csta.yourcompany.com
SIP/2.0

SIP-Call-ID: 19c4aacda45340a3b58eb9d602b2c073

SIP-CSeq: 1 INVITE

Data: user="johns@dvm1.csta.yourcompany.com"
```

Solution: Change the static routing rule or change the SIP URI for the user (in this example there was no routing rule for `dvm1.csta.yourcompany.com`).

6.2.Host Authorization not Configured or Incorrect

Symptom: The message flow in the logfile is similar to:

```
Direction: incoming

Peer: 10.0.10.5:26535 // this is the ip:port of the CSTA server

Message-Type: response

Start-Line: SIP/2.0 200 OK
```



```
From: <sip:Admin@yourcompany.com>;tag=f2bbc17a0c;epid=0d60acc870
To: <sip:callcontrol@dvm1.csta.yourcompany.com>;tag=228669937.656000
CSeq: 1 INVITE
Call-ID: 39413069451248c484e113bf7ecf11d6
Via: SIP/2.0/TCP
[...] ... a lot of messages are shown here before ... [...]
LogType: diagnostic
Severity: warning
Text: Message was discarded by the application
Result-Code: 0xc3e93ec6 SIP_E_AUTH_CANNOT_CHALLENGE
SIP-Start-Line: SIP/2.0 200 OK
SIP-Call-ID: 39413069451248c484e113bf7ecf11d6
SIP-CSeq: 1 INVITE
Data: application="http://www.microsoft.com/LCS/UserServices"
$$end_record
```

Solution: Set the IP address of the CSTA Server host as trusted via the host authorization rules. (In this example there was a matching routing rule, but the host 10.0.10.5 was not added to the list of authorized hosts.)

6.3.Checkbox “Replace Tel-URI” Not Enabled

Symptom: The message flow in the log file is similar to:

```
Direction: outgoing
Peer: 10.0.10.5:26535
Message-Type: request
Start-Line: INVITE sip:callcontrol@dvm1.csta.yourcompany.com SIP/2.0
```



```
From: <sip:Admin@yourcompany.com>;tag=b5b968571c;epid=0d60acc870

To: <sip:callcontrol@dvm1.csta.yourcompany.com>

CSeq: 1 INVITE
```

Solution: Check the checkbox labeled **Replace Tel-URI** in the OCS static routing rules for the CSTA Gateway server in question. (In this example, the **Replace Tel-URI** checkbox was not set so the symbolic name `dvm1.csta.yourcompany.com` was not replaced by the IP-address of the receiving host.)

6.4.Invalid Device Identifier

Symptom: the message flow in the logfile is similar to:

```
Direction: outgoing

Peer: 10.0.10.5:26535

Message-Type: request

Start-Line: INFO sip:callcontrol@10.0.10.5:26535;transport=tcp SIP/
2.0

From: <sip:Admin@yourcompany.com>;tag=23e5b8a60a;epid=0d60acc870

To: <sip:callcontrol@dvm1.csta.yourcompany.com>;tag=228744379.126000

CSeq: 3 INFO

Call-ID: ddf44c3ae51d49989c7ff897041512c6

...

<MonitorStart xmlns="http://www.ecma-international.org/standards/
ecma-323/csta/ed3"><monitorObject><deviceObject>tel:+4930282561000</
deviceObject></monitorObject></MonitorStart>

[ ... stripped of some message noise here ... ]

Direction: incoming

Peer: 10.0.10.5:26535
```



```
Message-Type: response

Start-Line: SIP/2.0 200 OK

From: <sip:Admin@yourcompany.com>;tag=23e5b8a60a;epid=0d60acc870

To: <sip:callcontrol@dvm1.csta.yourcompany.com>;tag=228744379.126000

CSeq: 3 INFO

Call-ID: ddf44c3ae51d49989c7ff897041512c6

...

Message-Body: <?xml version="1.0" encoding="UTF-8"?
><CSTAErrorCode><operation>invalidDeviceID</operation></
CSTAErrorCode>
```

The problem illustrated in this example was caused when the phone number configured for a given user doesn't match the phone numbers available on the PBX.

Solution: First verify that all the entries for all DID ranges appear in the DialPlan.conf file, that all entries are correct, and that all entries are in the correct order. If the DialPlan.conf file is valid, the problem lies with the **Line URI** entered into a user's configuration. It's common to find typographical errors such as entering **+14085552561** instead of **+14085555261**.

Canonicalized "international" format numbers (i.e. +14085551234) are required for user telephone number entries used in conjunction with OCS.

In the case that an incorrect telephone number entry was made in Active Directory and then corrected, the data in Active Directory must be resynchronized with OCS. An Address book sync up can be forced by using the following command from *\program files\microsoft office communications server 2007\server\core*: **Run abserver -syncnow**. You will see an entry in the event log indicating that the import has completed. Microsoft Office Communicator will download the new phone number for the user in question when it is restarted.



6.5.Changing Routing and/or Host Authorization Has No Effect

Solution: At a minimum, the front end service of the OCS needs to be restarted after changing the routing or authorization settings:To do so

- Stop front end service (via contextual menu, as shown)
- Start front end service (via contextual menu, as shown)

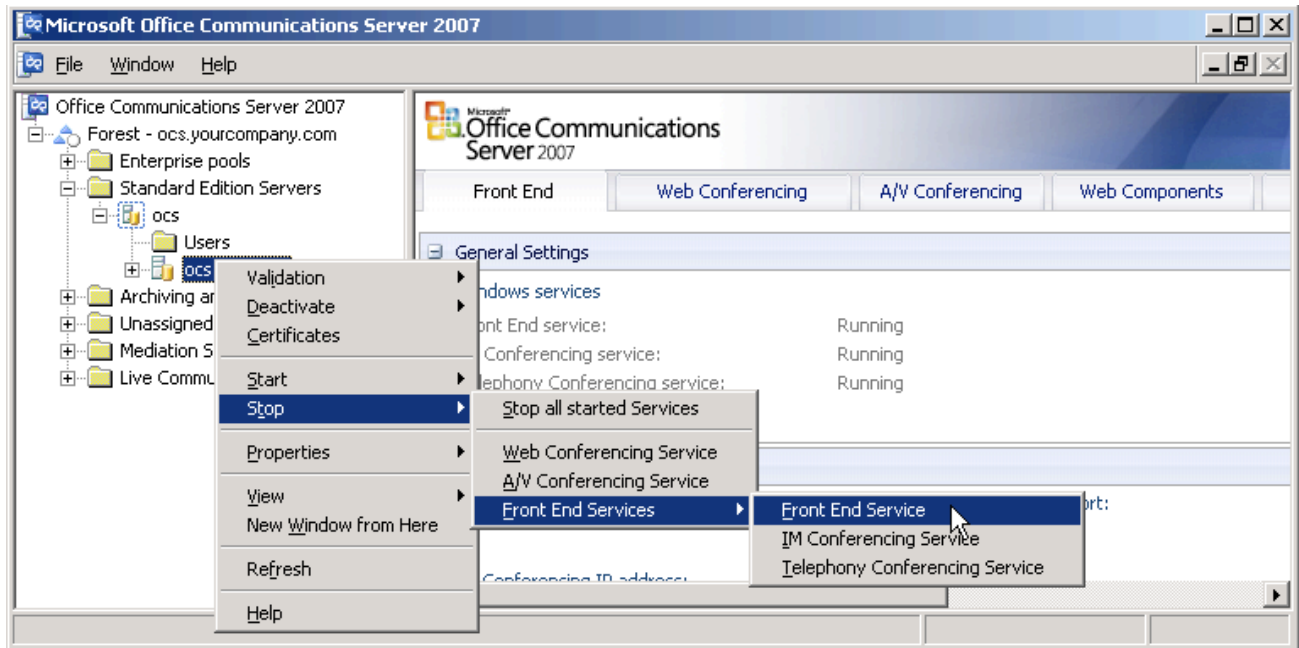


Figure 11 — Stopping the OCS front end service



7. TAPI Diagnostics

The ShoreWare CSTA Server takes advantage of ShoreTel PBX support for Microsoft Telephony Application Programming Interface (TAPI) to exchange call control service requests and events. As shown in Figure 10 on page 18, the ShoreTel-CSTAInk service connects to the ShoreTel TAPI service provider using the TAPI Service Provider Interface (TSPI).

The Telephony Management Service (TMS) application service (which runs on every ShoreWare server) connects the ShoreTel TAPI service provider to the rest of the ShoreTel distributed PBX. When TMS starts up, it creates a TAPI line device for each endpoint in the ShoreTel system. Access to these TAPI lines is provided through ShoreTel Remote TAPI Service Providers (RpcTsp.tsp and RpcTspX.tsp). The ShoreWare CSTA Server has access to these TAPI lines and receives new calls, call state information, and line device information from TMS via RpcTspX.tsp, thus allowing it to monitor all of the ShoreTel extensions.

To view the properties of the ShoreTel Remote TAPI Service Provider, open the **Phone and Modem Options** tab in the Windows Control Panel, then from the **Advanced** tab select **ShoreTel Remote TAPI Service Provider** and click on the **Configure** button.

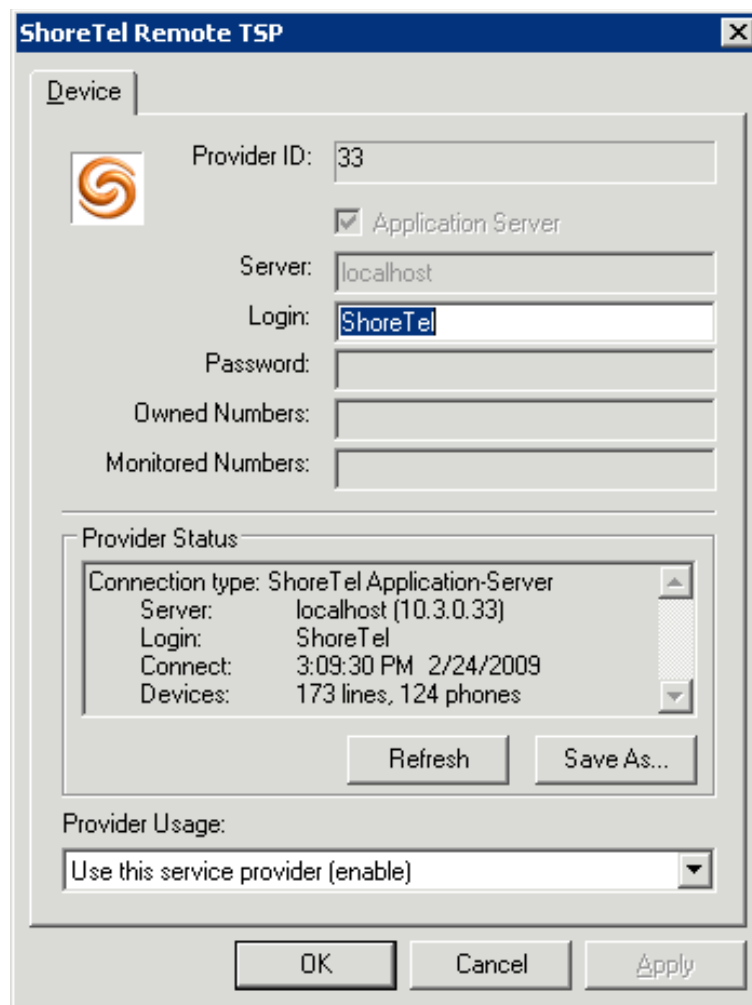


Figure 12—ShoreTel Remote TSP Properties



On a properly installed ShoreWare Remote Server (aka application server, DVS, DVM) the ShoreTel Remote TAPI Service Provider status should appear as shown in Figure 12. The **Application Server** and **Server** parameters are grayed out. The **Login** will be ShoreTel and no other parameters are defined. The **Provider Status** will display the server's localhost IP address, verify that this is the correct IP address. In addition, verify that the **Devices** status shows the appropriate number of lines and phones and that the **Provider Usage** is set to **enable**.

Several diagnostic steps involve observing the exchange of instructions and events between the PBX and the system running the Shoreware CSTA Server through this TAPI service provider. Diagnostics are done on the same host where the CSTA Server is running (the ShoreWare Remote Server).

Use Microsoft's **Dialer.exe** application (usually installed by default on the Windows 2003 server), to place a test call using one of the ShoreTel extensions. If it has never been used before the first step is to configure a TAPI line, see Figure 13 below.

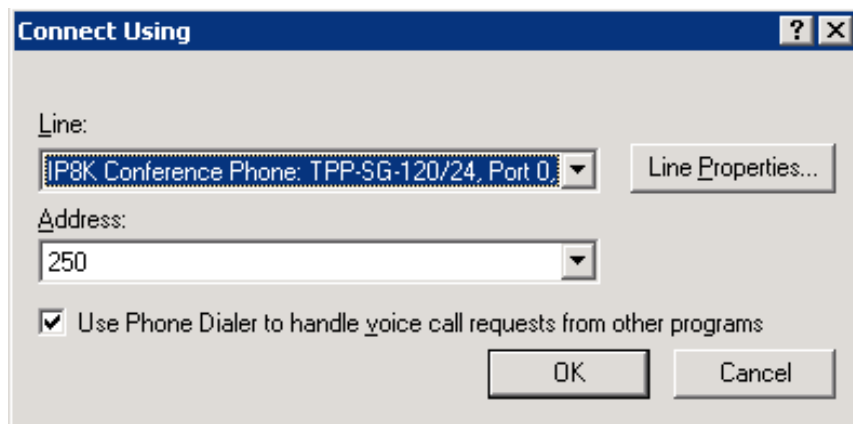


Figure 13—Configuring Dialer.exe

The **Line** is the name of a ShoreTel extension, along with the ShoreGear switch that is managing it and port number. The **Address** is the actual extension number. Select an appropriate Line and click on "OK". If the dialer application was utilized before and the **Connect Using** window is not displayed select **Tools** then **Connect Using....**

Note: If you cannot select the desired Line Address then there is probably something wrong with the ShoreTel system, TAPI, or a communication problem. Please refer to the appropriate ShoreTel Maintenance Guide or contact your service provider or ShoreTel's Technical Assistance Center.

Once you have configured the appropriate line, the **Phone Dialer** application will be active, see Figure 14 below

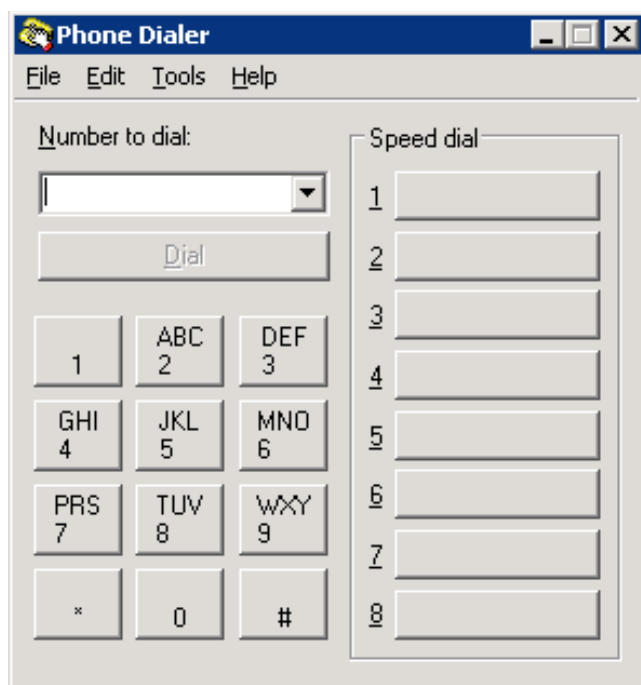


Figure 14—Phone Dialer window

To test dialing functionality:

1. To place a call to another extension enter a valid extension in the **Number to dial** field
2. Press the **Dial** button

Note: If you wish to place a call to an external number you must use an appropriate trunk access code followed by the full number (including country code). You cannot use a number like “+14085551212” here.

Note: If the TAPI phone dialer does not respond to the **Dial** command or shows an error then there is probably something wrong with the ShoreTel system, TAPI, or a communication problem. Please refer to the appropriate ShoreTel Maintenance Guide or contact your service provider or ShoreTel’s Technical Assistance Center.

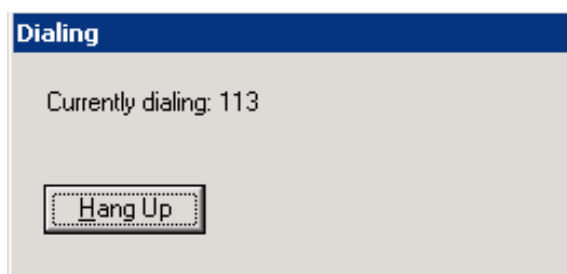


Figure 15—Dialing window

3. Press the **Hangup** button to end the call.
4. Close the Phone Dialer application by closing the application window or by selecting **Exit** from the **File** menu.



8. Logging

8.1.OCS Logging

The OCS logging facility has to be explicitly enabled using the OCS Logging Tool.

Use the panel tree on the left to navigate to the OCS domain node (DO NOT select the OCS host node). Right click on the OCS domain node to access the contextual menu and the select the **Logging Tool > New Debug Session** option.

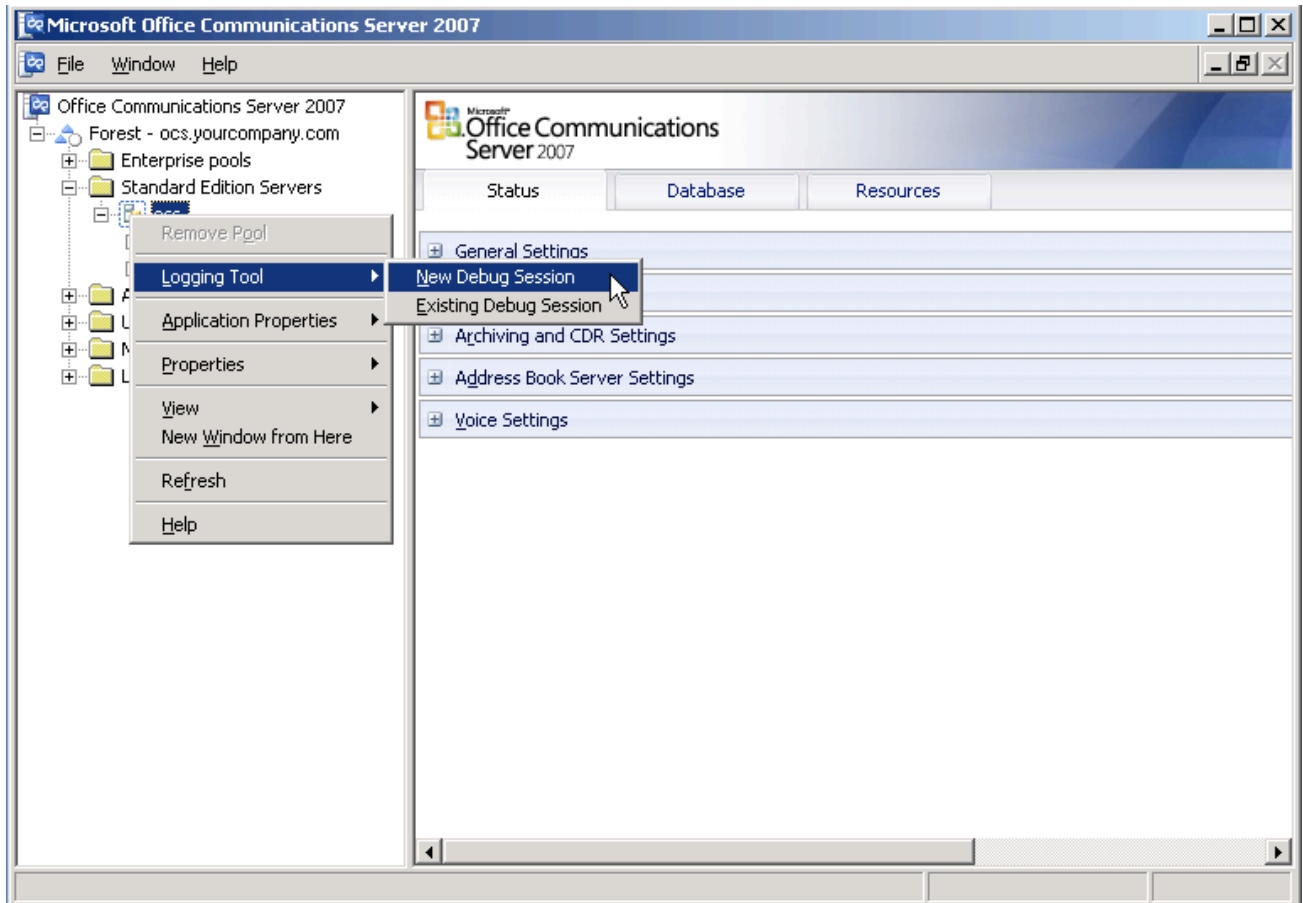


Figure 16—Starting a debug session

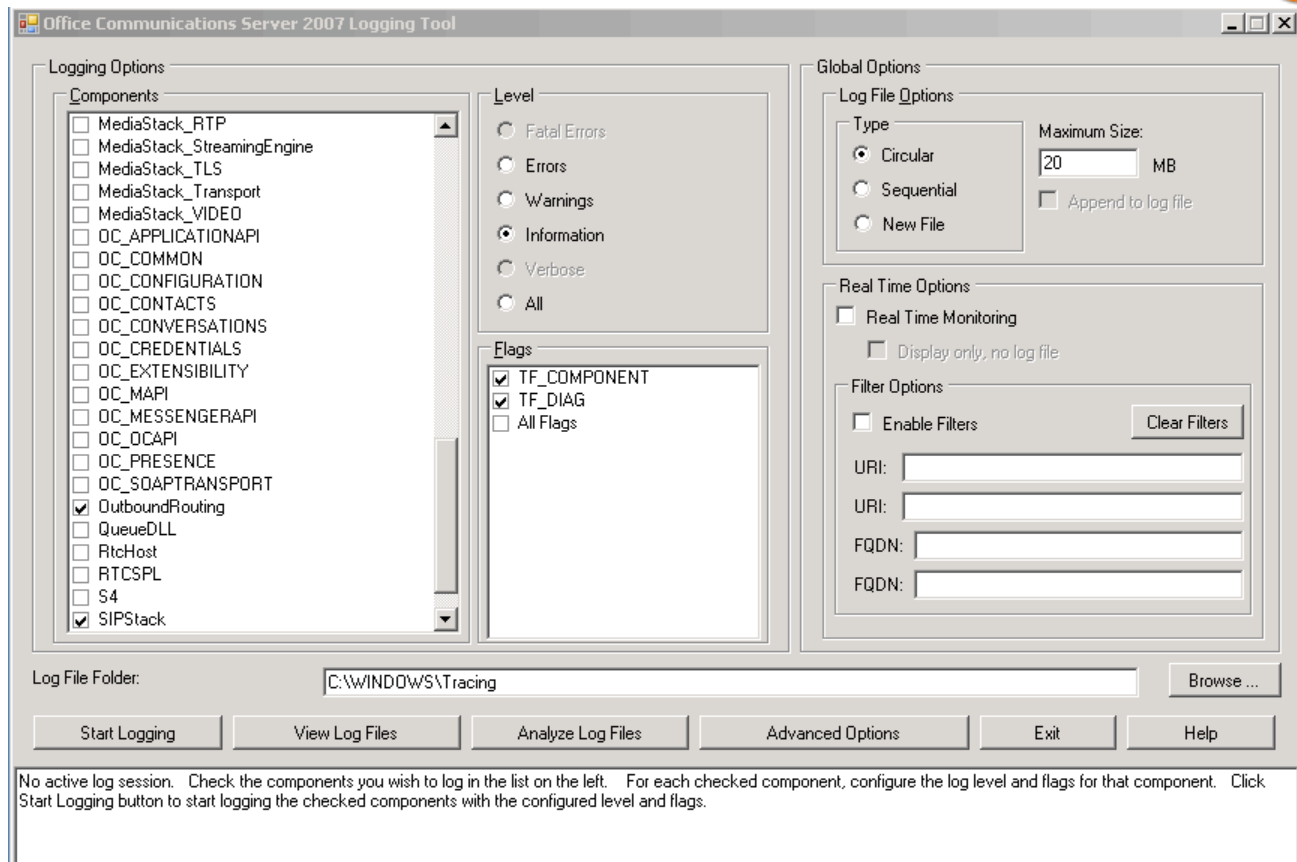


Figure 17—OCS 2007/OCS 2007R2 logging tool window

Inside the **Logging Tool** window the following components must be checked in order to troubleshoot OCS 2007/OCS 2007R2 telephony integration:

- Inbound Routing
- Outbound Routing
- SIPStack

The logging has to be explicitly started using the **Start Logging** button.

Note: The **Start Logging** button changes to a **Stop Logging** button when pressed.

After OCS logging has been started, sign-out/sign-in using the MOC client where the issue appears. Reproduce the failing scenario and then click the **Stop Logging** button.

The Log File Folder `C:\Windows\Tracing\` contains the log files in a binary format with filename suffix “.etl”.

To view the contents of these logs, click the **View Log Files** button in the **Logging Tool** window. The **View Log Files** window will then appear.

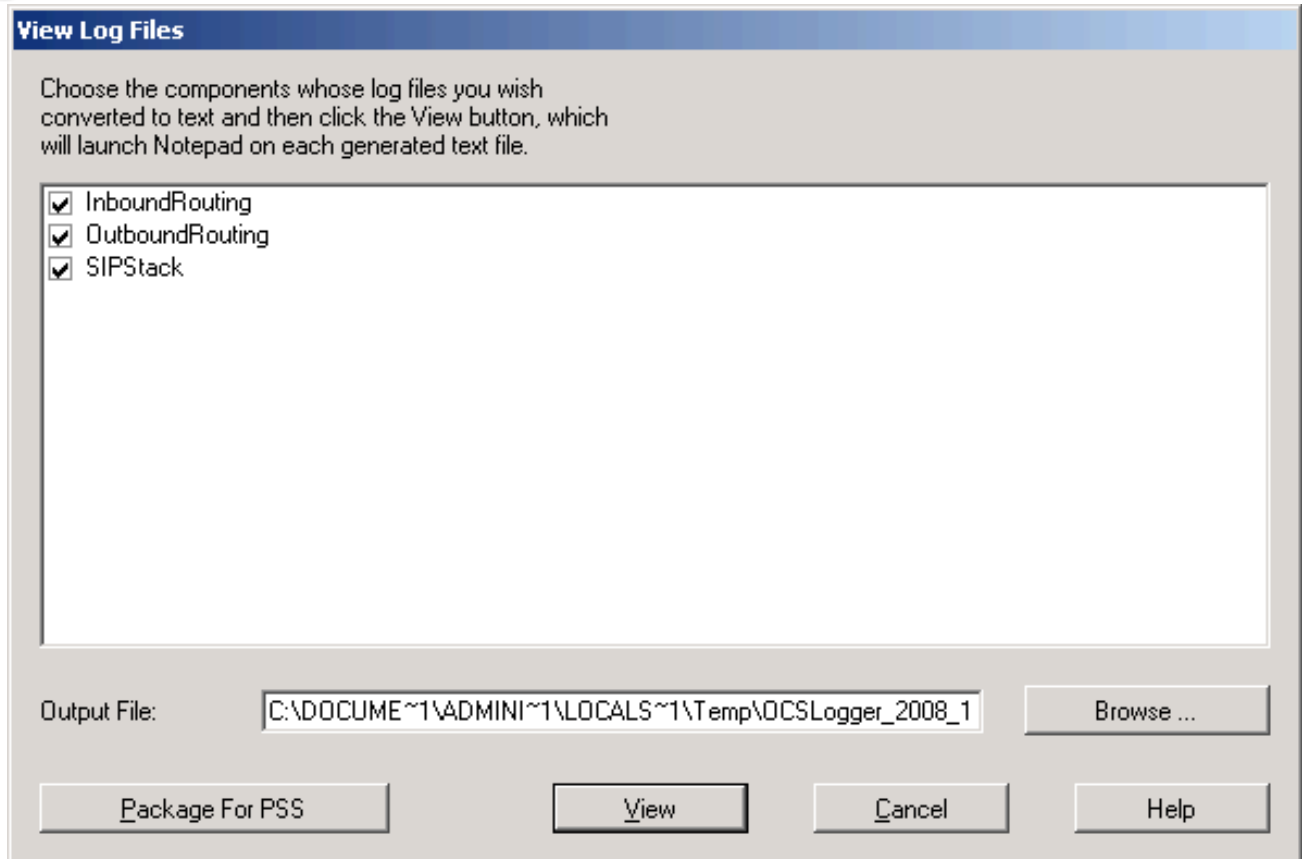


Figure 18—View Log Files window

Indicate which logging information is to be viewed using the checkboxes, specify the filename for the log to be read, and click the **View** button(1). This will convert the binary .etl files into a text file.



8.2.MOC Logging

MOC logging needs to be explicitly enabled as follows:

1. Open the **Options** window.

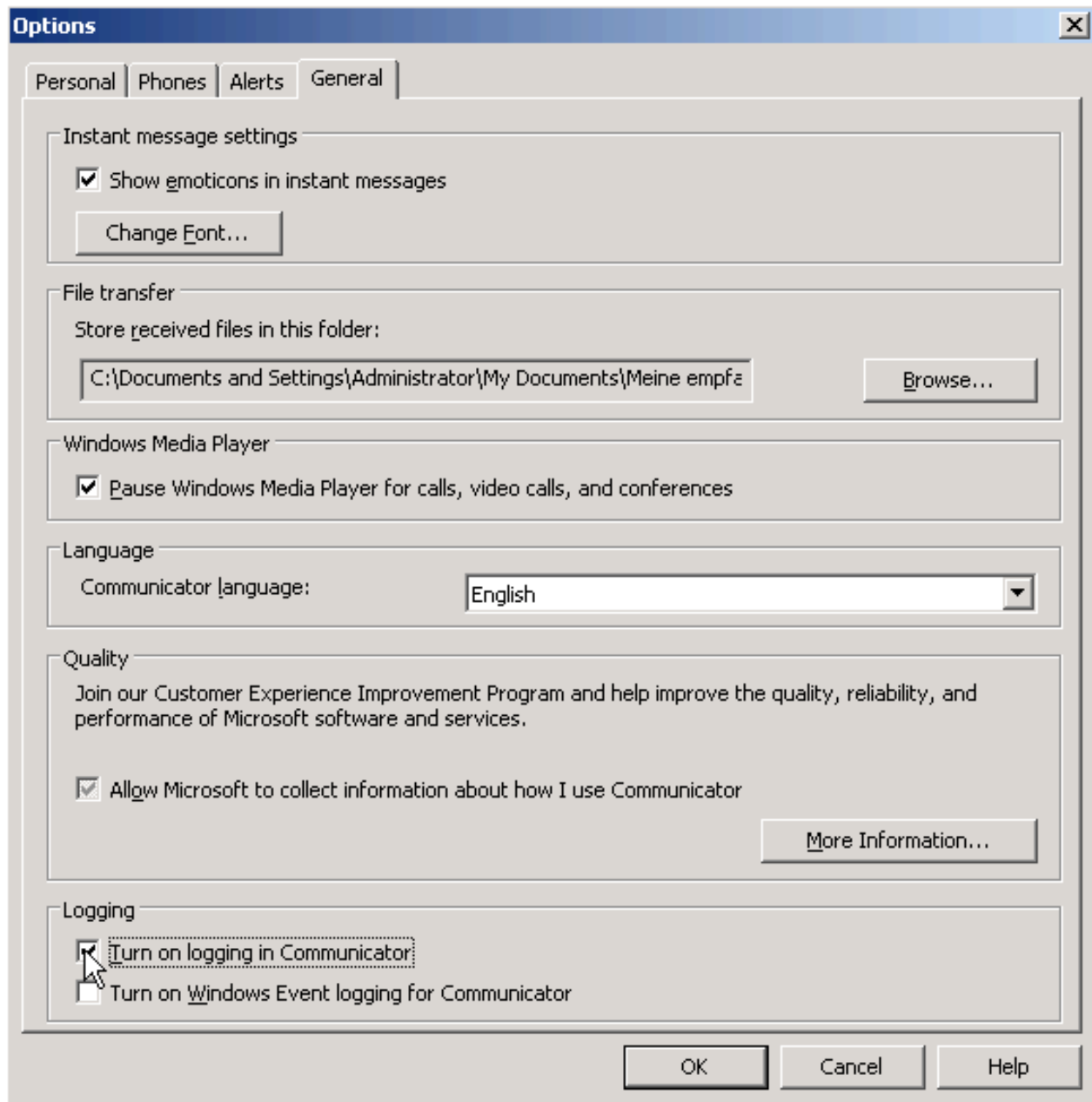


Figure 19—Turning on logging in MOC options

2. Select the **General** tab.
3. Check the **Turn on logging in Communicator** checkbox.
4. Re-login to the MOC.

A logfile is produced now. This logfile is stored as *C:\Documents and Settings\Administrator\Tracing\Communicator-uccp-0.uccplog*.



Example content of the *uucplog* file:

```
04/01/2008|13:55:14.156 2BC:940 INFO  :: SIP/2.0 504 Server time-out

Authentication-Info: Kerberos
rspauth="602306092A864886F71201020201011100FFFFFFFF6D7B7A0E04D10F53B
A62891944F98297", srand="BB548579", snum="10", opaque="0FDE32B5",
qop="auth", targetname="sip/ocs.yourcompany.com", realm="SIP
Communications Service"

From: <sip:Admin@yourcompany.com>;tag=f94e507772;epid=0d60acc870

To:
<sip:callcontrol@dvm1.csta.yourcompany.com>;tag=399A3A01D09653F93535
7E3180DFD0EB

Call-ID: 82fbde0b79474e5a9b6f09414d333191

CSeq: 1 INVITE

[...]
```

The above example illustrates a scenario where the remote CSTA server cannot be reached for one of the following reasons:

- Wrong static routing rules inside the OCS
- Network connectivity or firewall problem between OCS host and CSTA host
- CSTA Server is not up and running
- CSTA Server is configured with a wrong port number

Another example:

```
SIP/2.0

Via: SIP/2.0/TLS 192.168.46.10:2315

Max-Forwards: 70

From: <sip:Admin@yourcompany.com>;tag=ec0calc90f;epid=0d60acc870

To: <sip:callcontrol@dvm1.csta.yourcompany.com>;tag=228743379.844000

Call-ID: fdfd70fb2d7c4ec59e54c3bfde210eef

CSeq: 3 INFO
```



```
Route: <sip:callcontrol@10.0.10.5:26535;transport=tcp>

User-Agent: UCCP/2.0.6362.0 OC/2.0.6362.0 (Microsoft Office
Communicator)

Content-Disposition: signal;handling=required

Supported: timer

Proxy-Authorization: Kerberos gop="auth", realm="SIP Communications
Service", opaque="78BC6FFF", crand="2cc70834", cnum="10",
targetname="sip/ocs.yourcompany.com",
response="602306092a864886f71201020201011100ffffffff649328f1d401309f
3f2b43df34d2b241"

Content-Type: application/csta+xml

Content-Length: 203

<?xml version="1.0"?>

<MonitorStart xmlns="http://www.ecma-international.org/standards/
ecma-323/csta/ed3"><monitorObject><deviceObject>tel:
+49302852610001</deviceObject></monitorObject></MonitorStart>

[... more messages follow ...]

SIP/2.0 200 OK

Authentication-Info: Kerberos
rspauth="602306092A864886F71201020201011100FFFFFFFFF8776FF604DE1F1019
71163A976FAFB2F", srand="2D061025", snum="12", opaque="78BC6FFF",
gop="auth", targetname="sip/ocs.yourcompany.com", realm="SIP
Communications Service"

Via: SIP/2.0/TLS 192.168.46.10:2315;ms-received-port=2315;ms-
received-cid=400

From: <sip:Admin@yourcompany.com>;tag=ec0calc90f;epid=0d60acc870

To: <sip:callcontrol@dvm1.csta.yourcompany.com>;tag=228743379.844000

Call-ID: fdfd70fb2d7c4ec59e54c3bfde210eef

CSeq: 3 INFO

Contact: <sip:callcontrol@10.0.10.5:26535;transport=tcp>
```



```
Content-Disposition: signal;handling=required

Content-Type: application/csta+xml

Content-Length: 107

<?xml version="1.0" encoding="UTF-8"?
><CSTAErrorCode><operation>invalidDeviceID</operation></
CSTAErrorCode>
```

In this example the MOC sent a Monitor Start Request which indicates that there is no problem with routing, authorization, network connectivity, firewalls, etc.

The CSTA Server responds with a CSTAErrorCode of type invalidDeviceId to the MonitorStart request. This indicates that the wrong **Line URI** was specified in the OCS user specific phone settings.

Note: The Call-ID and the CSeq fields are very useful for tracing the flow of the SIP messages within other logfiles (e.g. from the OCS 2007/OCS 2007R2 server) because these attributes are used to correlate requests and responses.

8.3.CSTA Logging

The logfiles of the CSTA server are used primarily by ShoreTel to narrow down internal CSTA server issues.

However, ShoreTel might request that CSTA server logging be enabled to help diagnose a problem. Logging is enabled and configured via the CSTA server's configuration file *Default.conf* which in turn is normally located in the directory *C:\Program Files\Shoreline Communications\ShoreWare Server\CSTA\Config*. This file contains key/value pairs. The keys that control logging are as follows (default values are specified)

```
cstaLogEnabled = 0           # 0 -> off, 1 -> on
interfaceLogEnabled = 0      # 0 -> off, 1 -> on
debugLevel = 0               # 0 -> off, 9 -> max
```

By default, the logfiles generated by the CSTA Server are found in the subdirectory *C:\Program Files\Shoreline Communications\ShoreWare Server\CSTA\logs*.

Filename	Component	Content	Format
Sys.log	ShoreTel-CSTAsvr	Startup/Shutdown/licensing	
Error.log	ShoreTel-CSTAsvr	Internal processing	
CSTA.log	ShoreTel-CSTAsvr	Messages exchanged between ShoreTel-CSTAsvr and ShoreTel-CSTAcnr	hexdump
Interface.log	ShoreTel-CSTAsvr	Incoming Messages and outgoing messages correlated to the used protocol interfaces	hexdump
CSTAcnr.log	ShoreTel-CSTAcnr	Internal processing	
LinkTSP.log	ShoreTel-CSTAInk	Internal processing	